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April 26, 2007

MARKS & CLERK

Box 957
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Application No. : **2,466,918**
Owner : **SAMSUNG ELECTRONICS CO., LTD.**
Title : **METHOD AND APPARATUS FOR RECEIVING AND
DESHUFFLING SHUFFLED DATA IN A HIGH-RATE PACKET
DATA TELECOMMUNICATION SYSTEM**
Classification : **H04L 27/26 (2006.01)**
Your File No. : **16768-1-N.P.**
Examiner : **Sajith Bandaranayake**

YOU ARE HEREBY NOTIFIED OF A REQUISITION BY THE EXAMINER IN ACCORDANCE WITH SUBSECTION 30(2) OF THE *PATENT RULES*. IN ORDER TO AVOID ABANDONMENT UNDER PARAGRAPH 73(1)(A) OF THE *PATENT ACT*, A WRITTEN REPLY MUST BE RECEIVED WITHIN 6 MONTHS AFTER THE ABOVE DATE.

This application has been examined taking into account applicant's correspondence received in this office on May 10, 2006.

The number of claims in this application is 20.

A search of the prior art has revealed the following:

Reference Applied:

Patent Documents:

D1: WO 01/41313 A1

07 Jun 2001

Yonge III et al.

D1 discloses a transmit-configuration mechanism for configuring an FEC encoder including an associated interleaver for changing data channel characteristics. Channel information specifying a modulation mode and carriers capable of supporting the modulation mode for the data channel is received by a transmitting network node for use in a data transmission to a receiving network node. The received channel information is based on a prior data transmission to the receiving network node over the data channel. Configuration values are computed from the received channel information and an amount of data to be transmitted in a data transmission. The FEC encoder is configured to operate on the data transmission data amount according to the configuration information.

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The examiner has identified the following defects in the application:

Obviousness

D1 discloses the characteristics of a method and apparatus for receiving encoded and then shuffled data in a communication system (page 8 lines 16 - 22) supporting multi-level demodulation (page 15 line 32 - page 16 line 5) as set forth in claims 1 and 11:

- demodulating received data according to a predetermined demodulation scheme and outputting a modulation symbol having a predetermined number of code symbols (page 8 lines 2 - 7 and page 9 lines 13 - 23);
- deshuffling the code symbols in a deshuffling order corresponding to shuffling, the deshuffling order being determined considering the demodulation scheme (page 8 lines 19 - 32); and
- reading the deshuffled code symbols, decoding the code symbols at a predetermined code rate, and outputting a packet (figure 2 and page 8 lines 12 - 32).

Although D1 does not disclose a deshuffling memory device, this feature is considered an obvious design choice for a person skilled in the art and contains no inventive step that would distinguish claims 1 and 11 from D1 and the common general knowledge in the art. Thus the subject matter of claims 1 and 11 lacks an inventive step over the prior art and is deemed obvious.

Therefore, claims 1 and 11 do not comply with section 28.3 of the *Patent Act*. The subject matter of these claims would have been obvious on the claim date to a person skilled in the art or science to which they pertain, in view of D1 and the common general knowledge of any person skilled in the art of multi-level demodulation techniques.

In view of the foregoing defects, the applicant is requisitioned, under subsection 30(2) of the *Patent Rules*, to amend the application in order to comply with the *Patent Act* and the *Patent Rules* or to provide arguments as to why the application does comply.

Under section 34 of the *Patent Rules*, any amendment made in response to this requisition must be accompanied by a statement explaining the nature thereof, and how it corrects each of the above identified defects.

Sajith Bandaranayake
Patent Examiner
819-934-6754



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Sajith Bandaranayake
Patent Examiner
819-934-6754

WO0141313

Publication Title:

FORWARD ERROR CORRECTION WITH CHANNEL ADAPTATION

Abstract:

163e Abstract of WO0141313

A transmit-configuration mechanism (52) for configuring an FEC encoder (34) including an associated interleaver (84) for changing data channel characteristics. Channel information specifying a modulation mode and carriers capable of supporting the modulation mode for the data channel is received by a transmitting network node for use in a data transmission to a receiving network node. The received channel information is based on a prior data transmission to the receiving network node over the data channel. Configuration values are computed from the received channel information and an amount of data to be transmitted in a data transmission. The FEC encoder (34) is configured to operate on the data transmission data amount according to the configuration information.

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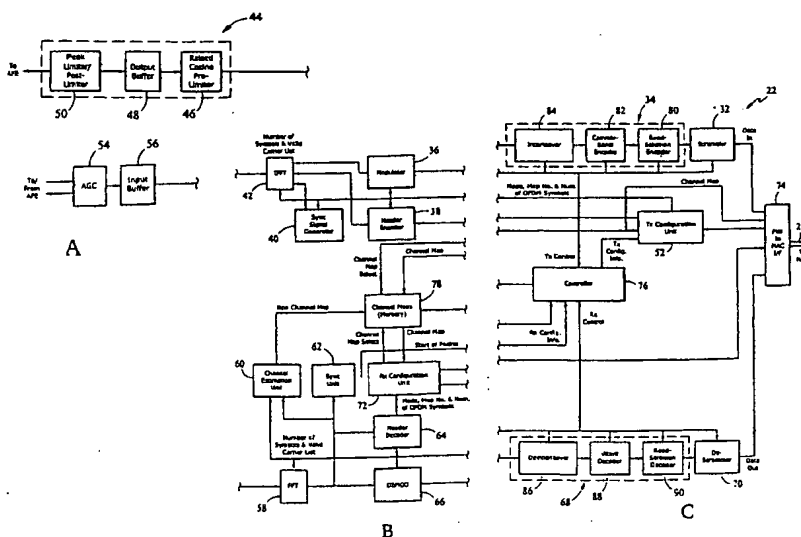
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(54) Title: FORWARD ERROR CORRECTION WITH CHANNEL ADAPTATION



(57) Abstract: A transmit-configuration mechanism (52) for configuring an FEC encoder (34) including an associated interleaver (84) for changing data channel characteristics. Channel information specifying a modulation mode and carriers capable of supporting the modulation mode for the data channel is received by a transmitting network node for use in a data transmission to a receiving network node. The received channel information is based on a prior data transmission to the receiving network node over the data channel. Configuration values are computed from the received channel information and an amount of data to be transmitted in a data transmission. The FEC encoder (34) is configured to operate on the data transmission data amount according to the configuration information.

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— Before the expiration of the time limit for amending the claims and to be republished in the event of receipt of amendments.

For two-letter codes and other abbreviations, refer to the "Guidance Notes on Codes and Abbreviations" appearing at the beginning of each regular issue of the PCT Gazette.

Forward Error Correction With Channel Adaptation

BACKGROUND OF THE INVENTION

The invention relates to OFDM data transmission systems.

5 OFDM is a spread spectrum technology wherein the available transmission channel bandwidth is subdivided into a number of discrete channels or carriers that are overlapping and orthogonal to each other. Data are transmitted in the form of symbols that have a predetermined duration and encompass some
10 number of carrier frequencies. The data transmitted over these OFDM symbol carriers may be encoded and modulated in amplitude and/or phase, using conventional schemes such as Binary Phase Shift Key (BPSK) or Quadrature Phase Shift Key (QPSK).

 A well known problem in the art of OFDM data
15 transmission systems is that of impulse noise, which can produce bursts of error on transmission channels, and delay spread, which often causes frequency selective fading. To address these problems, prior systems have utilized forward error correction (FEC) coding in conjunction with interleaving techniques. FEC
20 coding adds parity data that enables one or more errors in a code word to be detected and corrected. Interleaving reorders the code word bits in a block of code word data prior to transmission to achieve time and frequency diversity.

 Although the prior interleaving techniques can minimize
25 some of the effects of impulse noise and delay spread on OFDM data transmission, they cannot mitigate the impact of a combination of impulse noise and frequency nulls, which may result in lengthy noise events.

SUMMARY OF THE INVENTION

The present invention features a mechanism for adapting a forward error correction encoder (including an associated interleaver) to a channel.

5 In one aspect of the invention, a forward error correction (FEC) encoder is adapted to a channel over which data encoded by the FEC encoder and modulated onto OFDM symbols is to be transmitted in a data transmission to a receiving network node. Channel information specifying OFDM symbol block sizes
10 associated with the data transmission are received. The channel information is based on a prior data transmission to the receiving network node. FEC encoder configuration values are computed based on the received channel information. The FEC encoder is then configured to operate on the data according to
15 the configuration information.

Embodiments of the invention may include one or more of the following features.

20 The received channel information can be based on characteristics of the channel. It can include a modulation type and carriers capable of supporting the modulation type for the channel.

25 The OFDM symbol block sizes may include a fixed size and a variable size. The FEC encoder configuration values may be computed by determining from the channel information a number of OFDM symbol blocks of the fixed size and a number of remaining data bytes to be included in a last OFDM symbol block of the variable size, and computing a number of symbols for transmitting the remaining data bytes from the number of remaining data bytes, the channel information and a code block size associated with the
30 FEC encoder.

The FEC encoder can include an interleaver configured

to store the last variable size OFDM symbol block of encoded data based on the computed number of symbols and the channel information.

5 The FEC encoder can include a Reed-Solomon encoder. A maximum number of Reed-Solomon bytes in the last variable size OFDM symbol block can be computed. A Reed-Solomon block size can be computed from the maximum number of Reed-Solomon bytes in the last variable size block.

10 Among the advantages of the present invention are the following. In network having a transmitting network node communicating with a receiving network node over a data channel, and each of the nodes having a transmit portion and receive portion, the transmit portion in the transmitting network node can take advantage of information regarding recent channel
15 conditions as reflected in the most up-to-date channel map for the channel produced by the receive portion of the receiving network node based on a prior data transmission to the receiving network node. The FEC encoder configuration may be adjusted as necessary to accommodate variable OFDM symbol block sizes
20 dictated by the channel map updates.

Other features and advantages of the invention will be apparent from the following detailed description and from the claims.

25

BRIEF DESCRIPTION OF THE DRAWINGS

30

FIG. 1 is a data network of network nodes coupled to a power line transmission channel, each of the nodes in the data network including an end station (shown as a host computer), a media access control unit, a physical layer device and a analog front end unit.

FIG. 2 is a detailed block diagram of the physical

layer (PHY) unit (shown in FIG. 1) which includes, among other functional units, an FEC encoder (having as internal functional blocks an RS encoder, a convolutional encoder and an interleaver) on the transmit path, transmit (Tx)/ receive (Rx) configuration units, a channel maps memory and a channel estimation unit which operate collectively to configure the PHY unit for channel adaptation.

FIG. 3A is an illustration of the interleaver of FIG. 2 as written with encoded data by the convolutional encoder (also of FIG. 2).

FIG. 3B is an illustration of the interleaver as read by the modulator shown in FIG. 2.

FIG. 4 is an illustration of the arrangement of OFDM symbols stored in a single RAM of the interleaver for BPSK and QPSK modulations.

FIG. 5 is a flow diagram of the transmit (Tx) variable length OFDM symbol block computation for interleaver configuration as performed by the Tx configuration unit of FIG. 2.

FIG. 6 is a flow diagram of the process of determining the number of symbols to send in a last, variable size OFDM symbol.

FIG. 7 is a flow diagram of the transmit (Tx) variable length OFDM symbol block computation for Reed-Solomon block size modification as performed by the Tx configuration unit of FIG. 2.

FIG. 8 is a block diagram of a channel estimation unit (shown in FIG. 2), the channel estimation unit including a phase noise estimator for performing average phase noise computation for carriers and symbols in a received OFDM symbol block and a channel map generator.

FIGS. 9A and 9B is an illustration of phase noise computation for BPSK modulation and QPSK modulation,

respectively.

FIG. 10 is an illustration of the accumulation portion of phase noise averaging performed across both carriers and symbols.

5 FIG. 11 is a flow diagram of a channel map generation process which uses the phase noise to identify good carriers associated with BPSK and QPSK, and to select a modulation type.

DETAILED DESCRIPTION

10 Referring to FIG. 1, a network 10 includes network nodes 12a, 12b, ... 12k coupled to a data transmission medium shown as a power line (PL) 14. During a communication between at least two of the network nodes 12 over the data transmission medium, a first network node (for example, 12a) serves as a
15 transmitting network node and at least one second network node (for example, 12b) serves as a receiving network node. Each network node 12 includes an end station or device 16, e.g., a host computer (as shown), cable modem, etc.. The network node 12 further includes a media access control (MAC) unit 18 connected
20 to the end station 16 by a data interface 20, a physical layer (PHY) unit 22 connected to the MAC unit 18 by a MAC-to-PHY I/O bus 24 and an analog front-end (AFE) unit 26. The AFE unit 26 connects to the PHY unit 22 by separate AFE input lines 28a and output lines 28b, as well as connects to the power line 14 by an
25 AFE-to-PL interface 30.

Generally, the MAC and PHY units conform to the Open System Interconnect (OSI) Model's data link layer and the physical layer, respectively. The MAC unit 18 performs data encapsulation/decapsulation, as well as media access management
30 for transmit (Tx) and receive (Rx) functions. The PHY unit 22 performs transmit encoding and receive decoding, among other

functions, as described more fully below. The AFE unit 26 provides for attachment to the transmission medium, i.e., the power line 14. The MAC and AFE units may be implemented in a conventional manner and therefore will be discussed no further herein.

Referring to FIG. 2, the PHY unit 22 performs both Tx and Rx functions for a single node. To support Tx functions, the PHY unit 22 includes a scrambler 32, an FEC encoder 34, a modulator 36, a header encoder 38, a synchronization signal generator 40, an IFFT unit 42 and a post-IFFT block 44. The post-IFFT block 44 includes a raised cosine pre-limiter 46, an output buffer 48 and a peak limiter/post limiter 50. Also included are a transmit (Tx) configuration unit 52. To support Rx functions, the PHY unit 22 includes an automatic gain control (AGC) unit 54, an input buffer 56, an FFT unit 58, a channel estimation unit 60, a synchronization unit 62, a header/decoder 64, a demodulator 66, an FEC decoder 68, a descrambler 70, and receive (Rx) configuration unit 72. Included in the PHY unit 22 and shared by both the transmit and receive functions are a MAC interface 74, a PHY controller 76 and a channel maps memory 78.

During a data transmit process, data is received at the PHY-to-MAC interface (MAC interface) 74 over the PHY-to-MAC bus 24. The MAC interface provides the data to the scrambler 32, which ensures that the data as presented to the input of the FEC encoder 34 is substantially random in pattern. The FEC encoder 34 encodes the scrambled data pattern in a forward error correction code and subsequently interleaves the encoder data. Any known forward error correction code, for example, a Reed-Solomon, or, as shown, both a Reed-Solomon code and a convolution code, can be used for this purpose. The FEC encoder 34 therefore includes a Reed-Solomon (RS) encoder 80, a convolutional encoder 82 and an associated block interleaver 84. Although the block

interleaver is shown as part of the FEC encoder, it will be appreciated that the block interleaver 84 may be a separate component. The Reed-Solomon encoder 80 generates a code block from the scrambler output sequence using 256,239 RS encoding.

5 The convolutional encoder 82 receives as input the output from the RS encoder 80 and encodes that input with a standard rate equal to $1/2$ and a constraint length of $K=7$. When the last bit of data has been received at the convolutional encoder 82, it inserts 6 tail bits to flush out its internal shift register. As

10 known in the art, the convolutional encoder 82 is provided with a puncturing block for converting the convolutional encoder's output from a $1/2$ coding rate to some other coding rate, e.g., $3/4$. The modulator 36 reads the FEC encoded data from the interleaver 84 and encoded header information from the header

15 encoder 38, and modulates the encoded packet data onto carriers in OFDM symbols in accordance with conventional OFDM modulation techniques. Those modulation techniques may be coherent or differential. In the preferred embodiment, the modulation mode or type may be either Binary Phase Shift Keying with $1/2$ rate

20 coding (" $1/2$ BPSK"), Quadrature Phase Shift Keying with $1/2$ rate coding (" $1/2$ QPSK") or QPSK with $3/4$ rate coding (" $3/4$ QPSK"). The IFFT unit 42 receives input from the modulator 36, the header encoder 38 and synchronization signal generator 40, and provides processed packet data to the post IFFT block 44 described above,

25 which further processes the packet data before transferring it to the AFE unit 26 (from FIG. 1). Operational and implementation-specific details of the IFFT and post-IFFT block functional units 42, 44, respectively, are well known and, as they are not pertinent to an understanding of the present invention, will not

30 be discussed in any further detail.

The Tx configuration unit 52 receives information about the channel over which data is to be transmitted from the MAC

interface 74 and uses this information to select an appropriate channel map from the channel maps memory 78. The selected channel map specifies a transmission mode, as well as a modulation type (including an associated coding rate) and set of carriers to be used for the data transmission, and therefore specifies OFDM symbol block sizes (both fixed and variable) associated with the data transmission. An OFDM symbol block includes a plurality of symbols and may correspond to a packet or a portion thereof. The information read from the channel map is referred to herein as channel information. The Tx configuration unit 52 computes Tx configuration information from the channel information (i.e., channel map data). The Tx configuration information includes transmission mode, modulation type (including an associated FEC coding rate, as indicated above), number of symbols, number of bits per symbol, as well as number and size of Reed-Solomon blocks. The Tx configuration unit 52 provides the Tx configuration information to the PHY controller 76, which uses the information to control the configuration of the FEC encoder 34. More specifically, the controller 76 controls the interleaver configuration according to the carriers, number of bits per symbol and modulation specified by the Tx configuration unit 52. The controller 76 also enables modification of the RS block size based on the RS specific information computed by the Tx configuration unit 52. In addition to configuration control signals, the controller 76 also provides other conventional control signals to the FEC encoder 34, as well as the scrambler 32, the modulator 36, the synchronization signal generator 40 and the IFFT unit 42. The Tx configuration unit 52 also provides to the header encoder 38 the following header information: transmission mode, channel map number and number of OFDM symbols (to be transmitted) in a data packet.

During a data receive process, packets transmitted over the channel to the receiving network node 12b by the transmitting network node 12a are received at the PHY unit 22 from the AFE unit 26 by the AGC unit 54. The output of the AGC unit 54 is stored in the input buffer 56 prior to being processed by the FFT unit 58. The output of the FFT unit 58 is provided to the channel estimation unit 60, the synchronization unit 62, the header decoder 64 and the demodulator 66. More specifically, phase and amplitude values of the processed packet data are provided to the channel estimation unit 60, which produces a new channel map for use by the tx configuration unit 52 in the transmitting network node 12a during the next data transmission by that transmitting network node, as will be described. The Rx configuration unit 72 receives the mode, the channel map number and the number of OFDM symbols from the header decoder 64, retrieves the channel map specified by the map number provided by the header decoder 64, and provides Rx configuration information to the controller 76. The synchronization unit 62 provides a start-of-packet signal to the controller 76. In response to these inputs, the controller 76 provides configuration and control signals to the FEC decoder's functional units, which include a de-interleaver 86, a Viterbi decoder 88 and an RS decoder 90, and to the demodulator 66. More specifically, the PHY controller 76 specifies the appropriate number of rows and columns for the de-interleaver 86, the number of bits to be received by the Viterbi decoder 88 and the number and size of the Reed-Solomon blocks for decoding by the Reed-Solomon decoder 90. Additionally, it conveys the modulation type associated with the received packet data to the demodulator 66.

The demodulator 66 demodulates the OFDM symbols in the processed packet data received from the FFT unit 58 and converts phase angles of the packet data in each carrier of each symbol to

metric values. The demodulator 66 stores the metric values in a deinterleaver 86. The FEC decoder 68 reads the metric values from the deinterleaver 86 and uses the metric values for decoding purposes. The FEC decoder 68 corrects bit errors occurring
5 during transmission from the FEC encoder 34 (of a transmitting node) to the FEC decoder 68 and forwards the decoded data to the de-scrambler 70, which performs an operation that is the reverse of that which was performed by the scrambler 32. The output of the de-scrambler 70 is then provided to the MAC interface 74 for
10 transfer to the MAC unit 18 (and, ultimately, to an application of the host computer 16).

For purposes of simplification and clarity, details of the PHY unit's transmitter/receiver functional units which are known to those skilled in the art and not considered pertinent to
15 the understanding of the invention have been largely omitted herein.

Referring to FIGS. 3A and 3B, the interleaver 84 (as depicted during a write operation) is a row/column block interleaver memory of a fixed number M of columns 100 and a
20 variable number R of rows 110. In the preferred embodiment, $M=20$ and R is a value in the range of 28 to $2*N$, where N is equal to 84, the maximum number of usable carriers per symbol. The interleaver 84 stores an OFDM symbol block (or, simply, block) to be transmitted during a packet transmission. Each packet
25 includes one or more blocks. In the described embodiment, each block includes a maximum of 40 OFDM symbols. The interleaver 84 is implemented in a first 168x20 bit RAM 104 and a second 168x20 bit RAM 106. As illustrated by the arrows in FIG. 3A, data is stored by row, beginning with row 0. In the illustrated
30 embodiment, the interleaver 84 is capable of being operated in two different modes: a first ("standard") transmission mode and a second ("robust") transmission mode.

In the standard transmission mode, the interleaver 84 stores 40 OFDM symbol blocks associated with a packet, and is written in the following manner. For BPSK modulation type, only one of the two RAMs 104, 106 is used. The number of used rows is equal to two times the number of usable carriers per OFDM symbol. Twenty bits of encoded data are written into consecutive rows starting at row 0, as shown by the arrows. The least significant bit (LSB) of the twenty bit word is first-in-time encoded data. For QPSK modulation, both of the RAMs 104, 106 are used. After the first RAM 104 is filled from row 0 to row N (where $N=(2*\text{number of usable carriers})-1$), the second RAM 106 is filled starting at row 0.

The interleaver 84 is depicted during a read operation of a standard transmission in FIG. 3B. Referring to FIG. 3B, during a read operation for both BPSK and QPSK modulation types (or modes), the interleaver 84 is read by column with some amount of shifting to reorder bits. The modulator 36 reads by column starting at row 0, with each successive column read starting with the offset $p=8$ by adding eight to the previous column's start row. The row numbers (addresses) J are computed according to

$$J = (1+[(K-1)*p]) \bmod N \quad (1)$$

where K is the column number, p is an offset parameter (indicating the amount by which the column bits are shifted) and N is the total number of rows (or selectable memory locations). As an example, and as shown in the figure, if $K=2$, $p=8$, and $N=84$, the column read for the second column will begin at the ninth row (corresponding to carrier $N=8$). The LSB of the 20-bit word will be first-in-time modulated data. While the BPSK mode data is read from the first RAM 104 only, the QPSK mode data is read from both RAMS 104, 106 simultaneously. For each carrier, one bit is

read from each RAM using the same address to address both RAMs.

For the standard transmission mode (and both modulation modes), the number of usable carriers for the standard packet is determined from the channel map. The above technique provides data spreading in both time and frequency, allowing block errors in either domain to be spread far enough apart to allow correction by the FEC decoder 68.

The robust transmission mode interleaver varies from the standard packet interleaver in several ways. Its uses only the first 10 columns of the first RAM 104 and N rows (i.e., the number of usable carriers) to generate 10 OFDM symbols. As described in detail in co-pending U.S. application Ser. No. 09/377,131, in the name of Lawrence W. Yonge III, et al., incorporated herein by reference, the modulator 36 reads the interleaver 84 four consecutive times to create a 40 symbol packet containing four copies of the data. Alternatively, the robust transmission mode could be implemented to use some other number of columns, for example, 5 columns (for a 20 symbol block). Thus, the robust transmission mode interleaving process ensures that the data bit copies are not modulated onto adjacent carriers on a given symbol or neighboring symbols. Rather, they are spread uniformly in time and frequency so as to ensure successful decoding. While the redundant data need not be spread uniformly, it will be understood that the greater and more even the data copy spacing, the more robust the transmission is likely to be. In the illustrated embodiment, only the BPSK 1/2 rate coding modulation mode is used with robust transmission mode.

As it may be necessary or desirable to disable certain usable carriers, for example, it may be necessary to disable one or more of the 84 carriers so that the transmission does not interfere with the frequency bands of other RF services, the interleaver shift mechanism is adjustable for different numbers

of usable carriers. If the number of usable carriers is 83, for instance, the 1/4 offset requires a 20 row shift instead of the 21 row shift used for all 84 carriers and the shift mechanism will be adjusted accordingly.

5 Preferably, because the robust transmission mode has an additional level of data redundancy and therefore can only transmit a fraction of the amount of data that may be sent using the standard transission mode, it has a lower data rate than the standard transmission mode. Consequently, its use may be limited
10 to certain communications environments that require a high degree of reliability. For example, the robust mode may be particularly well-suited for use in broadcast transmission modes, in which the transmitting network node cannot adapt to each and every receiving network node because each of those nodes has a
15 different channel and those channels may have frequency nulls in different parts of the frequency band. Another use would be to establish initial contact between nodes which communicate over a power line. During such an initial set-up, a transmitting network node does not know which channel connects it to the
20 receiving network node and thus will transmit in a mode that the receiving network node can hear. However, the transmitting network node may not want to always transmit in the robust mode, as the robust mode may use too high a percentage of the channel. Thus, the transmitting network node's goal is to migrate to the
25 highest data rate as soon as possible so that other network nodes can use the channel. The transitting network node won't know what that data rate is until it has established initial communications with the receiver.

30 The interleaver control circuitry for controlling the mechanics of the reads and writes is well-known and therefore omitted from the description. Such control circuitry may be included in the convolutional encoder 82 and modulator 36, as is

assumed in the illustrated embodiment, in the interleaver 84 itself, or may reside in a separate control unit.

As indicated above, the FEC encoder 34 and the modulator 36 are controlled by the controller 76 to operate according to a selected combination of modulation mode type (including coding rate) and transmission mode (standard BPSK or QPSK, or BPSK robust mode), and selected set of usable carriers. The controller 76 provides such control in response to input from the Tx configuration unit 52, which reads an appropriate channel map from the channel map memory 78. As indicated above, the channel map defines for the channel each data transmission the mode type (robust mode, or standard modes BPSK or QPSK, as well as the coding rates for BPSK/QPSK) and those of the available carriers to be modulated with data (i.e., the usable carriers). Thus, it will be recognized that the number of blocks in a given packet transmission and the size of the last block in the packet sequence is determined by channel map, which may be frequently updated for changing channel conditions, as described below.

Referring to FIG. 4, an arrangement of OFDM symbols within a single one of the RAMs 104, 106 of the interleaver 84 as it is used for standard transmission mode 108 is shown. Each of the twenty columns 100 stores two OFDM symbols 110 (for a total of 40 OFDM symbols per block), with rows 0 to N-1 corresponding to the first symbol and rows N to 2N-1 corresponding to the second symbol. Thus, Column 0 corresponds to symbols 0 and 1, Column 1 corresponds to symbols 2 and 3, Column 2 corresponds to symbols 4 and 5, and so forth.

Referring to FIG. 5, a process of computing a variable block length (for a variable size block) 120 as performed by the Tx configuration unit 52 is shown. The unit 52 receives the channel information (with mode, code rate and usable carriers) in the form of a channel map (step 122). From the channel

information, the unit computes a maximum number of bytes per 40-symbol block (step 124). Once the maximum number of bytes has been determined, the unit 52 determines the total number of fixed-size (i.e., 40 symbols) blocks associated with the data transmission by dividing the total number of data bytes to be transmitted by the computed maximum number of bytes (per block) and rounding off the resulting quotient to the next lowest whole number (step 126). The remaining number of bytes is thus computed by subtracting the product of the maximum number of bytes per block and the number of blocks from the total number of data bytes (to be transmitted) (step 128). The unit 52 then computes the number of symbols to be included in the last, variable size block for the remaining number of bytes (step 130).

Referring to FIG. 6, the symbols number computation 130 is shown. First, the remainder is divided by the known maximum number of information bytes in an RS block (i.e., 239) to give a number of RS blocks (step 140). A block size for the RS block (including 16 bytes of parity) is then computed as the larger of two values, the minimum RS block size of 39 bytes or (the remainder divided the number of RS blocks) plus 16 (step 142). The total number of symbols upon which bits to be modulated on symbols after coding can then be determined by dividing a total number of bits in the RS block (that is, [the RS block size*number of RS blocks*8]+6) by the total number of bits per symbol (the number of usable carriers times the code rate) to give the number of symbols (step 144). As the number of OFDM symbols must be an even number due to the interleaver implementation, the result is converted to an even number as follows: the result is multiplied by 1/2 and rounded up to the next whole number, which is multiplied by 2 to produce an even number of symbols (step 146).

Once the number of symbols in the variable size last

block is known, the Tx configuration unit 52 provides the computed number of symbols and the number of bits per symbol (based on the number of usable carriers, the modulation type and code rate), collectively referred to as interleaver configuration information, to the controller 76. The controller 76 uses the interleaver configuration information to configure the interleaver 84 for a number of columns based on the number of symbols and for a number of rows based on the number of bits per symbol.

Referring to FIG. 7, the Tx configuration unit 52 also produces RS encoding configuration information 150 so that the RS block size can be modified for variable block length. The unit 52 subtracts 6 bits from the total number of bits to be encoded (i.e., the number of symbols times the number of bits per symbol times the code rate), divides that value by 8 and rounds down the resulting bytes value to the next lowest whole number to give the maximum number of RS bytes to be put on symbols of the last packet (step 152). The unit then computes the number of RS blocks in the block by dividing the sum of the maximum number of RS bytes and the maximum number of information bytes plus one by the maximum RS block size plus 1 and rounds down (step 156). The RS block size is computed as the smaller of 255 (i.e., the maximum RS block size) or rounded down quotient of the maximum number of RS bytes divided by the number of RS blocks (step 158). The unit uses these values to compute the maximum number of data bytes as the product of the RS block size minus the parity and the number of RS blocks (step 160).

It should be noted that, to compute the maximum number of RS bytes for robust mode blocks, the unit uses a predetermined number, e.g., 10 (as indicated above) or 5, defined for the robust transmission mode, as the number of symbols, a BPSK code rate of 1/2 and a number of bits per symbol equal to the number

of usable carriers. The maximum number of data bytes is computed by subtracting eight from the maximum number of RS bytes.

Returning to FIG. 1, the demodulator 66 demodulates the modulated carriers using a scheme appropriate to the modulation technique used by the modulator 36. The demodulator 66 produces from a phase angle for each bit of the transmitted carrier data a 3-bit soft decision ("bit metric) value from 0 to 7 that represents a probability of a "0" or a "1" bit, with 7 representing a "1" and 0 representing a "0". A phase difference is determined using the following equation:

$$D_o = \text{mod}((2\pi + \theta_k) - \Psi_k, 2\pi) \quad (2)$$

where D_o is the k_{th} carrier phase difference, θ_k is the current symbol's k_{th} carrier phase, Ψ_k is the previous symbol's k_{th} carrier phase and 2π radians is the maximum phase value. The phase difference D_o is converted to a value of 0-127 points ($2\pi=128$). D_o is then offset by an amount, depending on the modulation type, to allow for a single soft decision conversion.

The de-interleaver 86 (of FIG. 2) receives the 3-bit soft decision value for each data bit. All 3-bit soft decision values are de-interleaved (i.e., stored in the de-interleaver) as a group. The method for writing the interleaver 84 applies to reading the de-interleaver 86 and the method of reading the interleaver 84 applies to writing the de-interleaver 86. The write operation uses the reverse algorithm of that applied to the interleaver during a read operation.

Referring to FIG. 8, the channel estimation unit 60 includes a rectangular to polar converter 160, a controller 162, referred to herein as a phase noise estimator 162, and a channel map generator 164. The rectangular to polar converter 62 receives complex numbers from the FFT unit 58 and converts the

complex numbers to polar to give a corresponding phase angle sample. The output of the converter 160 is provided to the phase noise estimator 162, which produces average carrier phase noise values PN_c and symbol phase noise values PN_s , in the manner described in above-referenced application. The phase noise estimator 62 monitors the phases and amplitudes of each carrier in each OFDM symbol as they are received from the FFT unit 58. The phase noise estimator 162 computes the phase noise associated with each carrier and each symbol by performing phase noise estimation, phase noise estimation accumulation and averaging. The phase noise estimation can be performed for either BPSK or QPSK, that is, whichever modulator type was used by the modulator. For BPSK, a binary 1 causes the transmission of a zero phase and binary 0 the transmission of a Π phase. Thus, in the case of BPSK, which sends only the two states (corresponding to "1" and "0"), the phase noise estimator measures how far the samples are from the expected 1 and 0 values.

The constellation plot for the sample may be represented in binary form, with 0 to 2π radians being represented as 0 to 127 (or 0 to 255) points. For a given sample X, the phase noise computation estimator 162 computes a phase noise estimation for the carrier frequency of that sample. It then computes an average of the computed phase noise values for each carrier frequency as well as each symbol. The average may be expressed as

$$PN_{avg} = (\sum Y1) / (\text{total number of samples}) \quad (3)$$

where $Y1 = |Y - (\Pi/2)|$ and $Y = \text{mod}[X + (\Pi/2); \Pi]$. The value Y1 is the phase noise and is expressed in terms of number of points from the ideal expected modulation values, which in the case of BPSK are zero or Π , the zero or Π states being indicative of no noise.

The phase angle is represented in binary form as a phase number between 0 and 127 (or 0 and 255). The phase noise computation estimator 162 creates a modulus of a phase number y , e.g. 64 (or 32), adds $y/2$ points, and finds $X+(y/2) \bmod y$. It then subtracts $y/2$ so that the result is always a value between $-y/2$ and $+y/2$. Once the phase noise estimator 162 obtains the absolute value of that value, the result lies in the first quadrant (0 to $y/2$) of the constellation.

An exemplary phase noise calculation for BPSK is illustrated in FIG. 9A. In the constellation plot of the example shown, 2π radians is represented as a binary value corresponding to 128 points. For a sample having a phase number of 80, the calculation adds 32 to give a sum of 112 and computes $(112 \bmod 64)$. Thus, referring to Eq. (3), Y equals 48 and $Y1$ is equal to the absolute value of $(48-32)$, or 16 points.

A similar phase noise computation may be performed for QPSK, which uses four states (or phases) spaced $\pi/2$ apart. An exemplary QPSK phase noise estimation is illustrated in FIG. 9B.

The phase noise average of Eq. (3) may be computed for phase noise as a function of the carrier, the symbol, or both. To compute the carrier phase noise average, PN_c , the phase noise estimator accumulates carrier values for a given carrier for all of the symbols and divides by the total number of symbols. In the described embodiment, the total number of symbols in an OFDM packet is 40. Thus, PN_c is the average phase noise for a carrier for an entire block of data. Additionally, for a symbol phase noise average, PN_s , the phase noise across all carriers in a symbol is accumulated and divided by total number of carriers (i.e., 84). The PN_s value provides an indication of how carrier phase noise varies (relative to PN_c) from symbol to symbol. Thus, the combination provides a reasonable estimate of signal-to-noise (S/N) for a given carrier on a symbol-by-symbol basis.

Referring to FIG. 10, the accumulation (or summation) of phase noise values for a given carrier frequency over time and by symbol 170 is depicted. The phase noise values for each carrier 171 are accumulated by summing the phase noise values for each carrier over the forty OFDM symbols 172 to give a sum, $SUM(PNC(M))$ 174, where M is one of the usable carriers. Similarly, the phase noise values are accumulated for each OFDM symbol 172 by summing the phase noise values for all usable carriers 171 to give a sum, $SUM(PNS(N))$ 176, where N is one of symbols 1 through 40. The total number of symbol accumulations or sums is therefore 40. Any carriers not used by the transmission are excluded from the summation.

As discussed in the above-referenced co-pending application, the phase estimator/controller 162 uses above described phase noise computation to weight the four robust transmission mode copies differently (i.e., the copy samples with less phase noise are weighted more heavily than the copy samples with more phase noise) and recombines the weighted copies prior to decoding.

Referring back to FIG. 8, the channel map generator 164 receives the PN_s and PN_c values and uses those values to select a modulation type/code rate and associated "good carriers" based on the channel conditions as reflected in the average phase noise values. The selected mode/rate and associated carriers, once obtained, are defined in a channel map, which is stored in the channel map memory 78 (of FIG. 2) and also sent to the MAC (via the PHY-to-MAC interface 74) for transfer to the transmitting network node. Once stored in the channel map memory 78, the channel map is available to and can be accessed by the Rx configuration unit 72 (FIG. 2) for decoding during the next data transmission over the channel to which the channel map corresponds.

The channel map generator 164 uses six types of thresholds to assess the conditions of the channel. Included are two symbol thresholds, a first symbol threshold THDS1 and a second symbol threshold THDS2. Also included among the five
5 threshold types are three carrier thresholds, THDM1, THDM2 and THDM3, corresponding to modulation types BPSK with 1/2 rate error coding ("1/2 BPSK"), QPSK with 1/2 rate error coding ("1/2 QPSK") and QPSK with 3/4 rate error coding ("3/4 QPSK"), respectively. Also employed is a jammer detection threshold for detecting
10 jammer signals on a carrier by carrier basis. Therefore, these six thresholds, in conjunction with the average phase noise values (for symbols and/or carriers) and carrier amplitudes, are used to generate information (for each modulation type) indicative of noise events that occurred in the data packet.

15 Referring to FIG. 11, the channel map generation process 164 is shown. The process receives from the phase noise estimator 162 the PN_s and PN_c values for a block of data being demodulated and decoded (step 180). The process then determines a first impulse noise count PNS_{c1} , the total sum (or count) of PN_s
20 values exceeding THDS1, and a second impulse noise count PNS_{c2} , the total sum (or count) of PN_s values exceeding THDS2 (step 182). The PNS_{c1} and PNS_{c2} values provide an indication of impulse noise at the levels corresponding to THDS1 and THDS2, respectively. That is, the counts provide an indication of the
25 amount of impulse noise on the channel or, more specifically, how often a noise event of sufficient magnitude causes a particular symbol's noise level to exceed one or both of the thresholds THDS1 and THDS2. Preferably, the thresholds THDS1 and THDS2 are adjusted dynamically on a block-by-block basis using an average
30 of the PN_s values for each block. Thus, for a block having an average PN_s of some value X, THDS1 may be set to a first threshold value $X + y1$ and THDS2 set to a second threshold value $X + y2$,

where y_1 and y_2 are constants and $y_1 < y_2$ so that THDS1 and THDS2 correspond to respective different levels of impulse noise.

Still referring to FIG. 11, the process finds the total number (or count) of carriers for which the PN_c values which are below each of the carrier thresholds THDM1, THDM2 and THDM3, as well as the jammer detection threshold (step 184). It computes a first count, $BPSK_c$, as the number of carriers having a PN_c value less than the 1/2 BPSK threshold, THDM1, and an amplitude less than the jammer detection threshold. It computes a second count, $QPSK_{c1}$, as the number or count of carriers whose PN_c value is less than the 1/2 QPSK threshold, THDM2, and whose amplitude is less than the jammer detection threshold. A third count, $QPSK_{c2}$, corresponding to the number or count of carriers having a PN_c value that is less than the 3/4 QPSK threshold, THDM3, as well as an amplitude below the jammer detection threshold, is also computed.

Therefore, an accumulator counts the number of times the average phase noise is better than the threshold level. That is, the PN_c is compared to THDS1 and THDS2, and PN_c is compared to THDM1, THDM2 and THDM3 thresholds. The PN_c counts are incremented only when the phase noise average is below the threshold and the amplitude value is below a jammer detect threshold for the same carrier.

As the PN_c for each of the carriers is compared to each of the three carrier thresholds, the generator produces an 84-bit "good carriers" channel map for each of the three modulation types 186. More specifically, for each of the 84 carriers (carriers 0-83), a bit is set in the channel map if the PN_c is below the threshold and the amplitude value is below the jammer detect threshold.

Additionally, the process computes average threshold margins MAR1, MAR2 and MAR3 for the good carriers represented by

each of the carrier counts $BPSK_c$, $QPSK_{c1}$ and $QPSK_{c2}$, respectively (step 188). During each carrier threshold comparison for each modulation type, if the carrier threshold noise requirement is exceeded (that is, PN_c is less than the threshold), the amount by which it is exceeded is accumulated. After all of the threshold comparisons for a given threshold have been performed, the accumulated number is divided by the number of times the threshold requirement was exceeded to give an average margin (i.e., amount below the threshold).

As the carrier phase noise averages PN_c are compared to the three thresholds and channel maps are generated, the process performs a test for each of the modulation types associated with the standard transmission to determine if certain criteria have been met for the good carriers indicated by their corresponding channel maps (step 190). The criteria for each modulation type is provided in Table 1 below.

Modulation Type	Criteria (*)
	(*) For each modulation type, THDS1= 13.0 (0 dB SNR) and THDS2 = 9.0 (4.5 dB SNR)
1/2 BPSK	<u>For THDM1 = 10.4</u> a) CNTM1 > 42 (greater than 42 good carriers, requires 3 dB SNR) b) MAR1 > 1.5 (margin > 1.5dB) c) CNTS1 < 8 (less than 8 corrupted symbols)
1/2 QPSK	<u>For THDM2 = 7.3</u> a) CNTM2 > 21 (greater than 21 good carriers, requires 6 dB SNR) b) MAR2 > 1.5 (margin > 1.5 dB) c) CNTS2 > 6 (less than 6 corrupted symbols)

3/4 QPSK	<p>For THDM3 = 5.0</p> <p>a) CNTM3 > 21 (greater than 21 good carriers, requires 9 dB SNR)</p> <p>b) MAR3 > 1.5 (margin 1.5-2 dB)</p> <p>c) CNTS2 = 0 (no impulse noise)</p> <p>d) Average P_{N_s} < 8 (5 dB SNR)</p>
----------	---

TABLE 1

The process thus determines if any one or more of the modulation types associated with the standard transmission mode meet the specified criteria (step 192). If the process determines that one or more of the modulation types associated with standard transmission mode meets the specified criteria, for each modulation type that meets the specified criteria (as shown in Table 1), the process computes the relative data rate based on the number of carriers (specified by the appropriate one of CNTM1, CNTM2, or CNTM3) and selects the modulation type with the highest data rate (step 194); however, the highest modulation rate must be faster than the next highest modulation rate by some predetermined amount (e.g., 10%) faster to be selected.

Otherwise, the process selects the next highest modulation rate. The "good carriers" channel map for the selected modulation type (and coding rate) is then stored in the channel maps memory (step 196). Using BPSK as an example, with 1 bit per carrier and a CNTM1=50 at a 1/2 rate error coding, the data rate is equal to $1/2 \times (50 \text{ bits/symbol time})$ or 25 bits/symbol time. Therefore, the relative data rate is 25. With 1/2 QPSK having the same count and 2 bits per carrier, the relative data rate is $1/2 \times (100) = 50$. For 3/4 rate QPSK with CNTM3=50, the relative data rate is $3/4 \times (100) = 75$. In this example, then, based on the relative rates of 25, 50 and 75 for 1/2 BPSK, 1/2 QPSK and 3/4 QPSK, respectively, the process selects 3/4 QPSK. If the criteria is

not satisfied for any of the three modulation types (at step 192), then the channel is too noisy to for the transmission rates of the standard transmission mode and the more reliable robust transmission mode is selected (step 198).

5 Once stored in the channel maps memory 78, the channel map is available to the Rx configuration unit 72 as well as the controller 76, which directs a copy to the PHY-to-MAC interface for transfer to the transmitting network node. The transmitting network node stores the channel map in its own channel maps
10 memory 78 for subsequent use by its Tx configuration unit 52 when preparing a next data transmission to the receiving network node over the channel to which the map corresponds. Alternatively, if the robust transmission mode is selected, the selection of the robust transmission mode is conveyed to the transmitting network
15 node.

Other Embodiments

It is to be understood that while the invention has been described in conjunction with the detailed description
20 thereof, the foregoing description is intended to illustrate and not limit the scope of the invention, which is defined by the scope of the appended claims. Other embodiments are within the scope of the following claims.

For example, the channel estimation unit could select
25 modulation mode/coding rates on a carrier by carrier basis. This "bit loading" technique would allow a different amount of modulation to be put on each carrier based on the signal-to-noise ratios. Although the above processes are described with respect to 1/2 and 3/4 coding rates, other coding rates (e.g., 7/8) may
30 be used. Also, other FEC coding and interleaving schemes may be used.

Alternatively, or in addition to the above, channel

13 the data according to the configuration information.

1 27. An apparatus for adapting a forward error correction
2 (FEC) encoder to a channel over which data encoded by the FEC
3 encoder and modulated onto OFDM symbols is to be transmitted in a
4 data transmission to a receiving network node, comprising:
5 means for receiving channel information specifying
6 symbol block sizes associated with the data transmission, the
7 received channel information being based on a prior data
8 transmission to the receiving network node;
9 means for computing FEC encoder configuration values
10 based on the received channel information; and
11 means for configuring the FEC encoder to operate on the
12 data according to the configuration information.

1/15

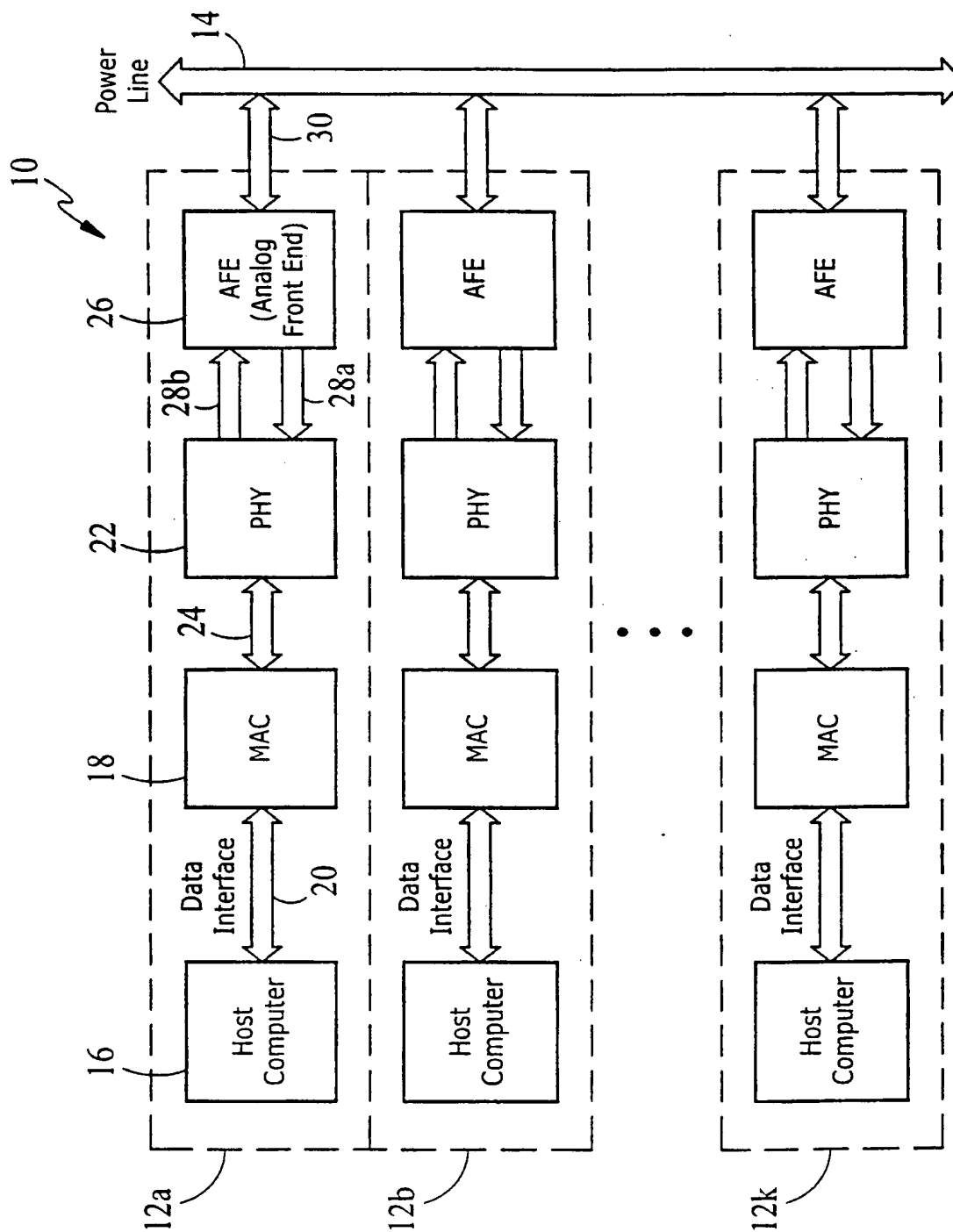


FIG. 1

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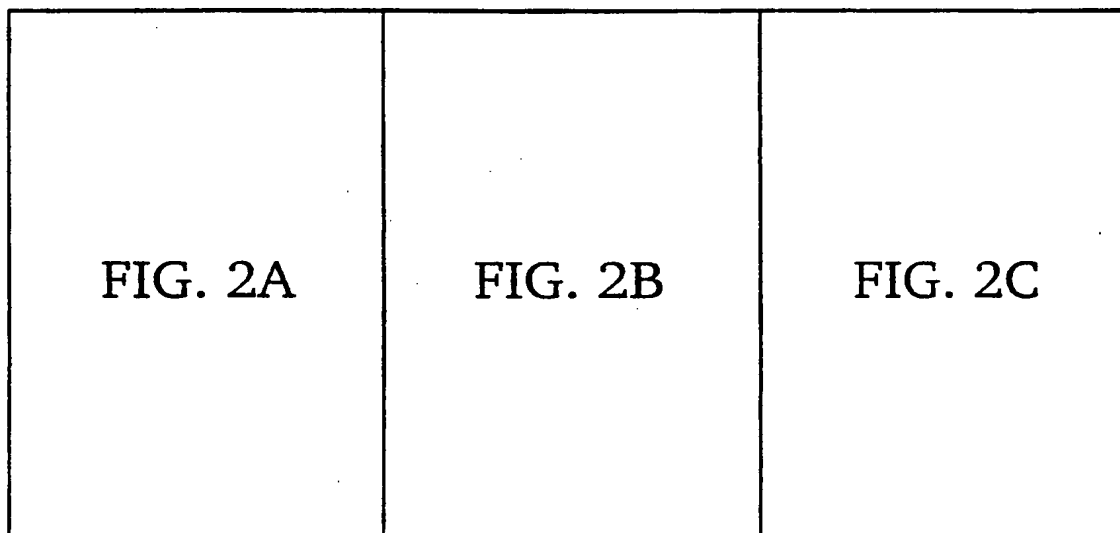


FIG. 2

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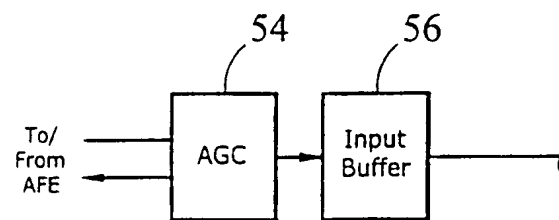
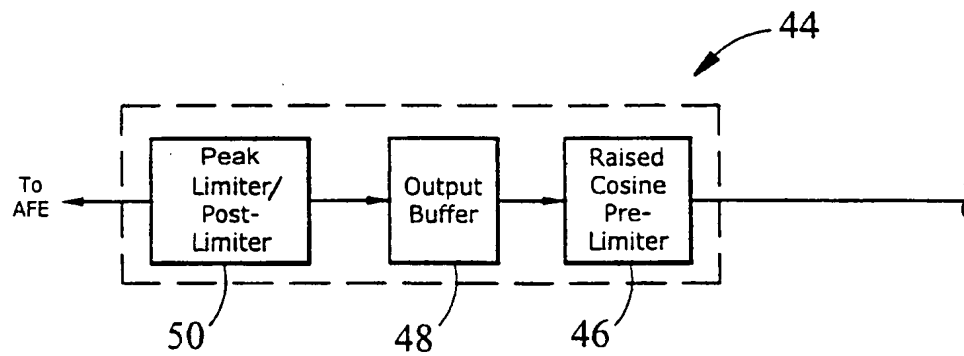


FIG. 2A
SUBSTITUTE SHEET (RULE 26)

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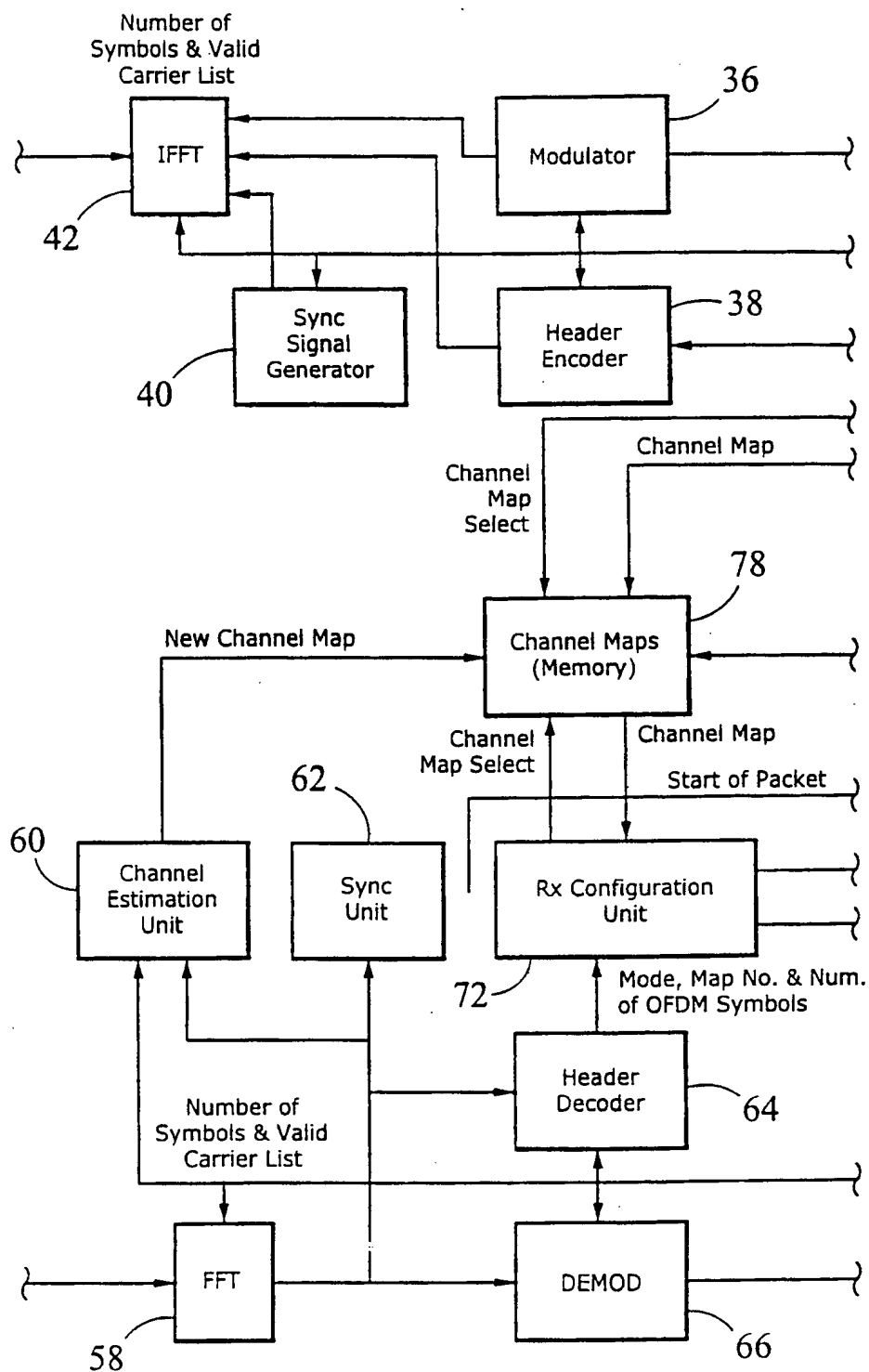


FIG. 2B

SUBSTITUTE SHEET (RULE 26)

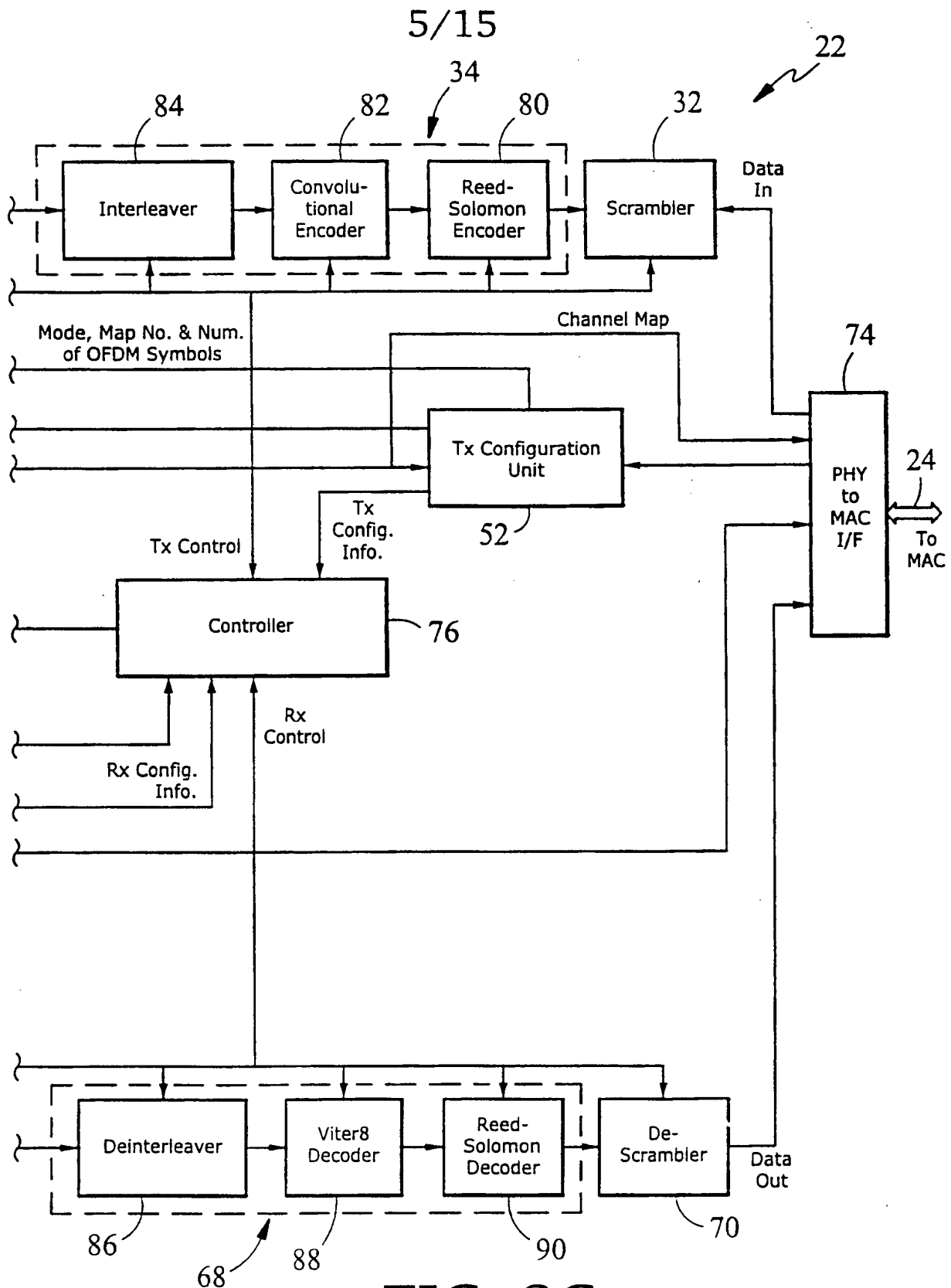


FIG. 2C
SUBSTITUTE SHEET (RULE 26)

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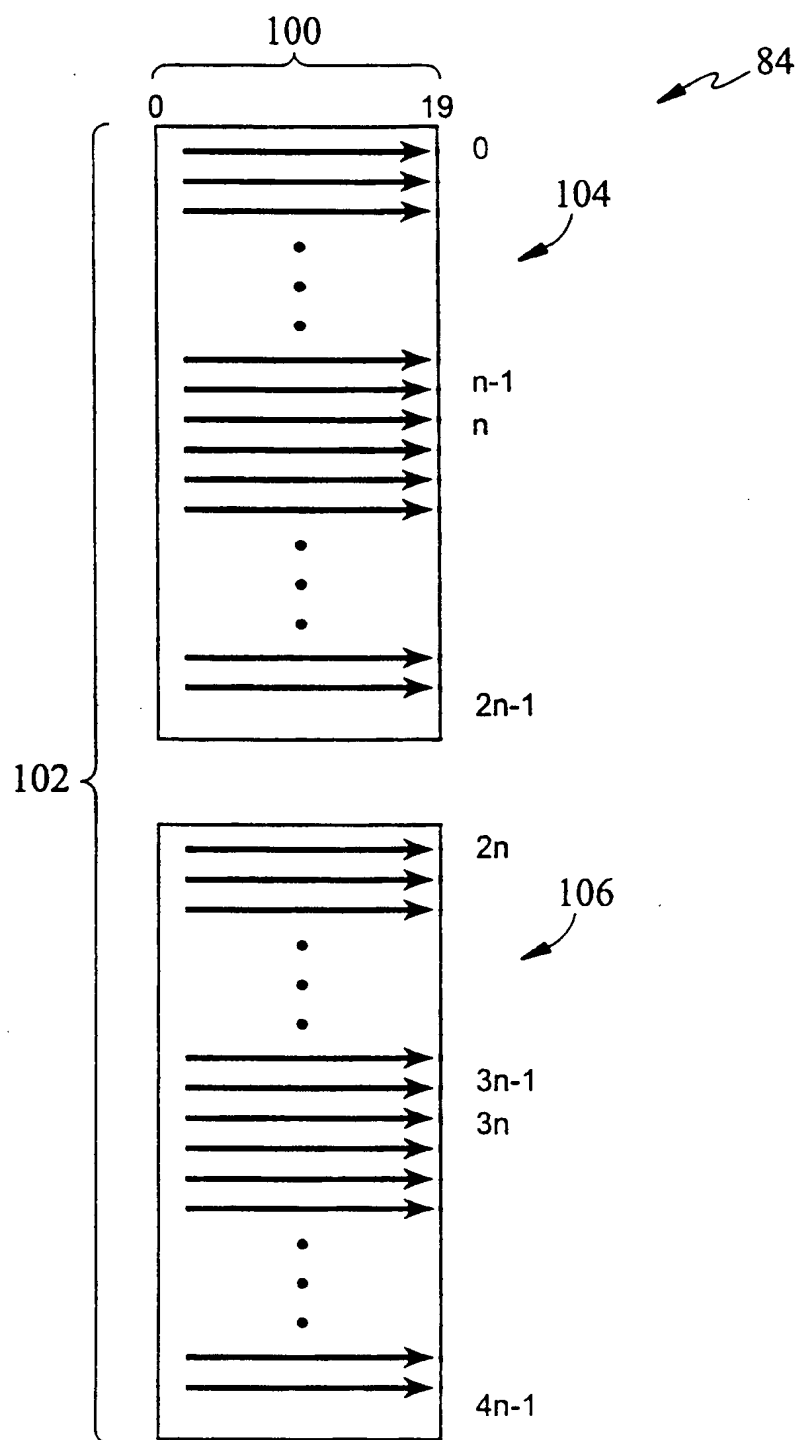


FIG. 3A

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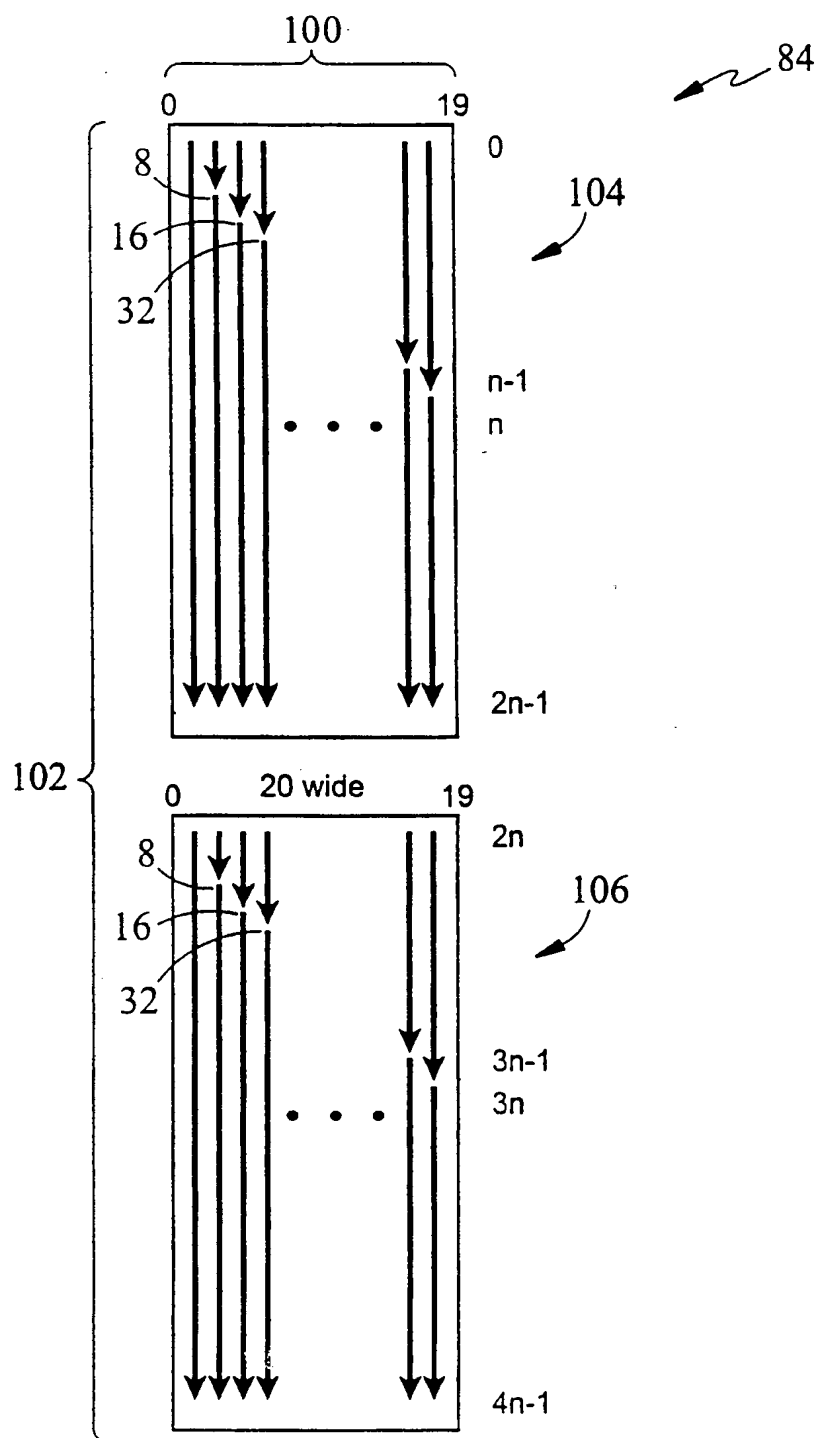


FIG. 3B

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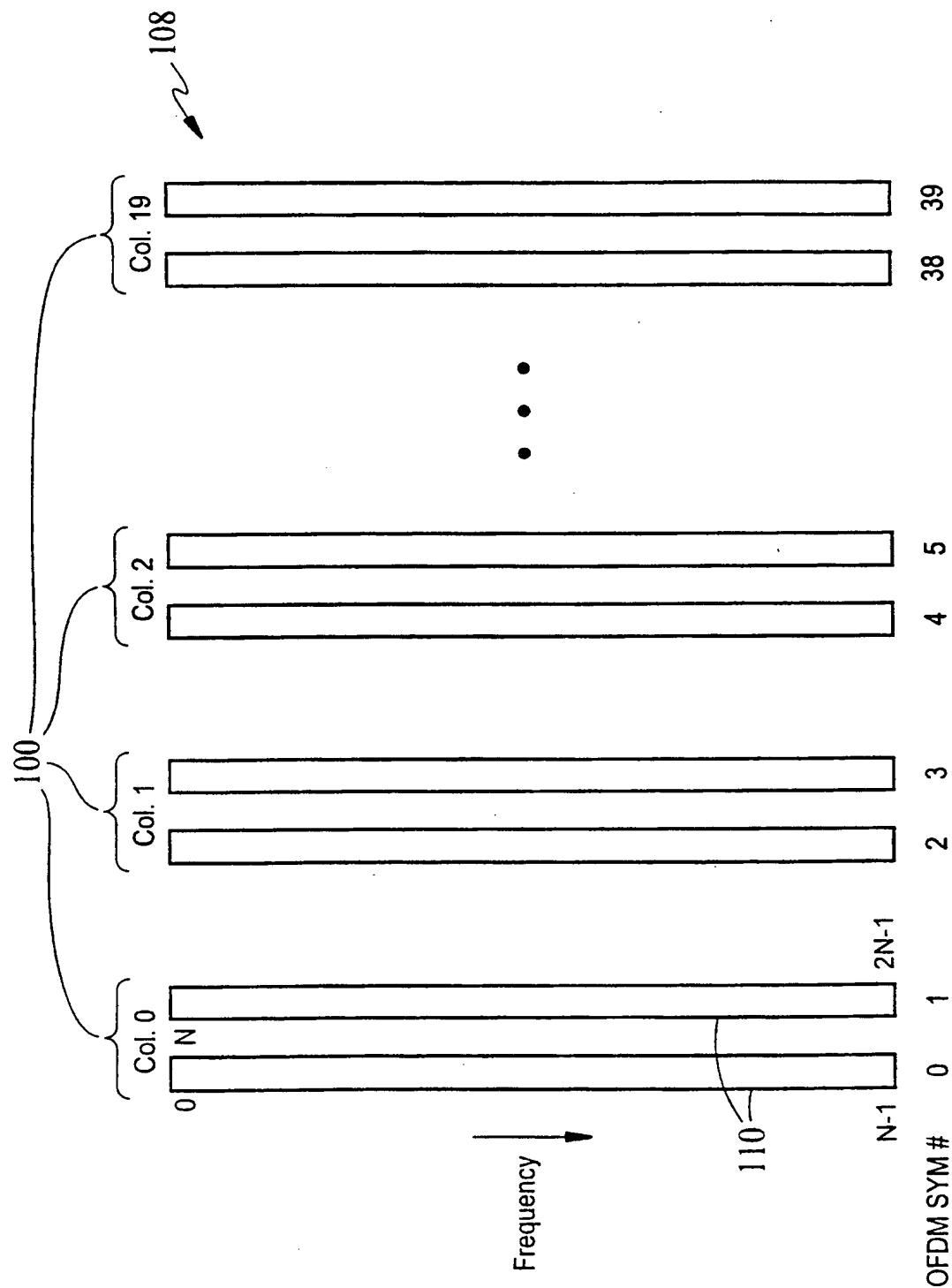


FIG. 4

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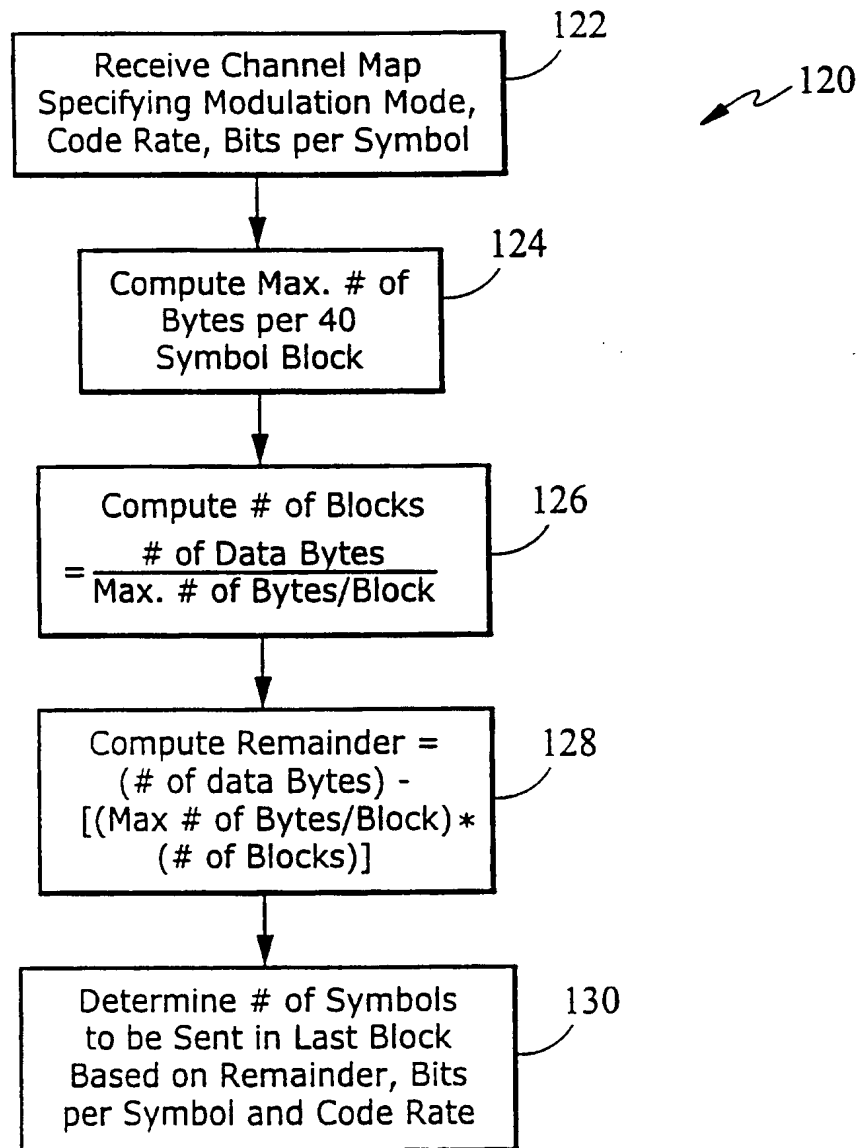


FIG. 5

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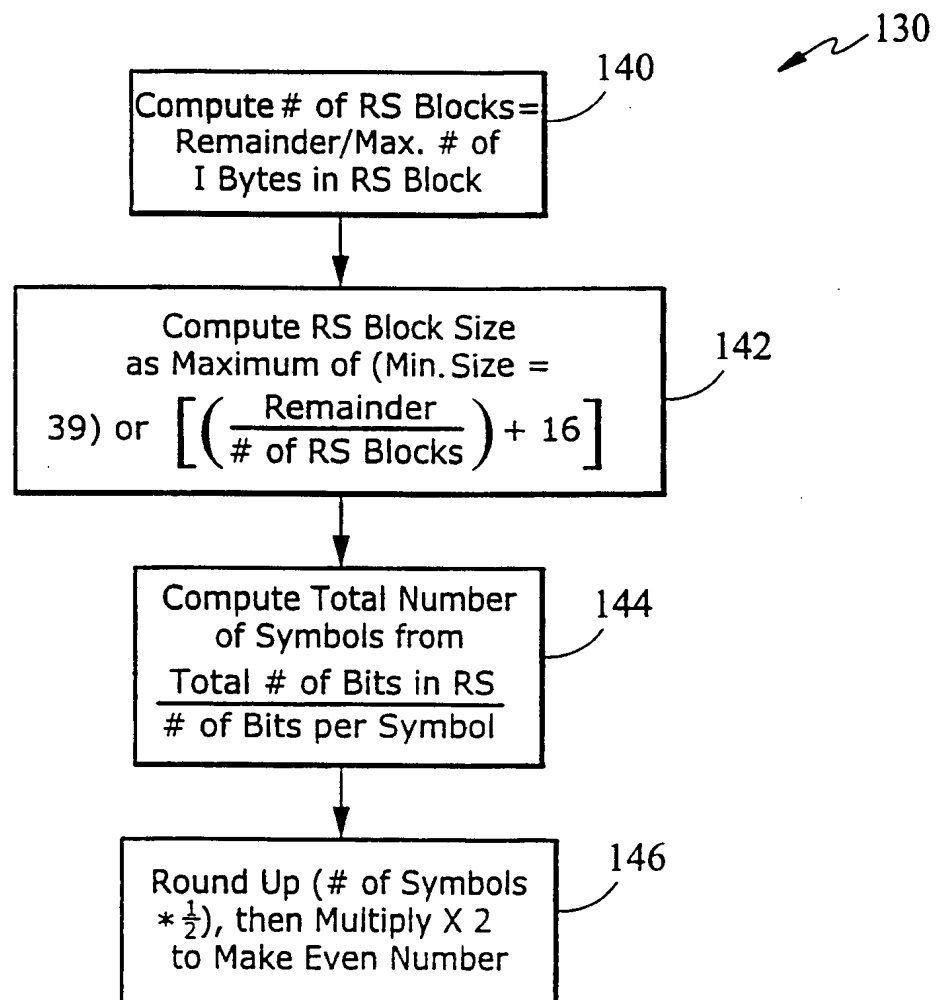


FIG. 6

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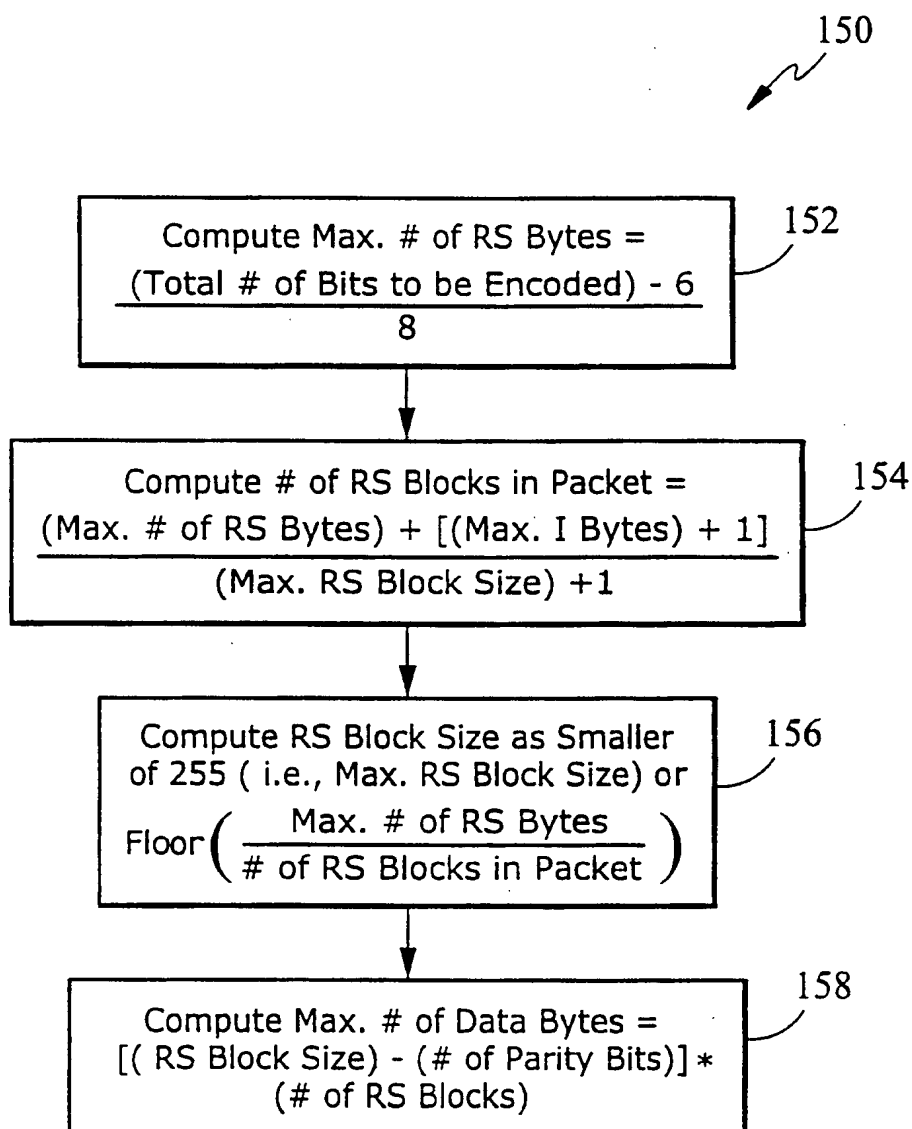


FIG. 7

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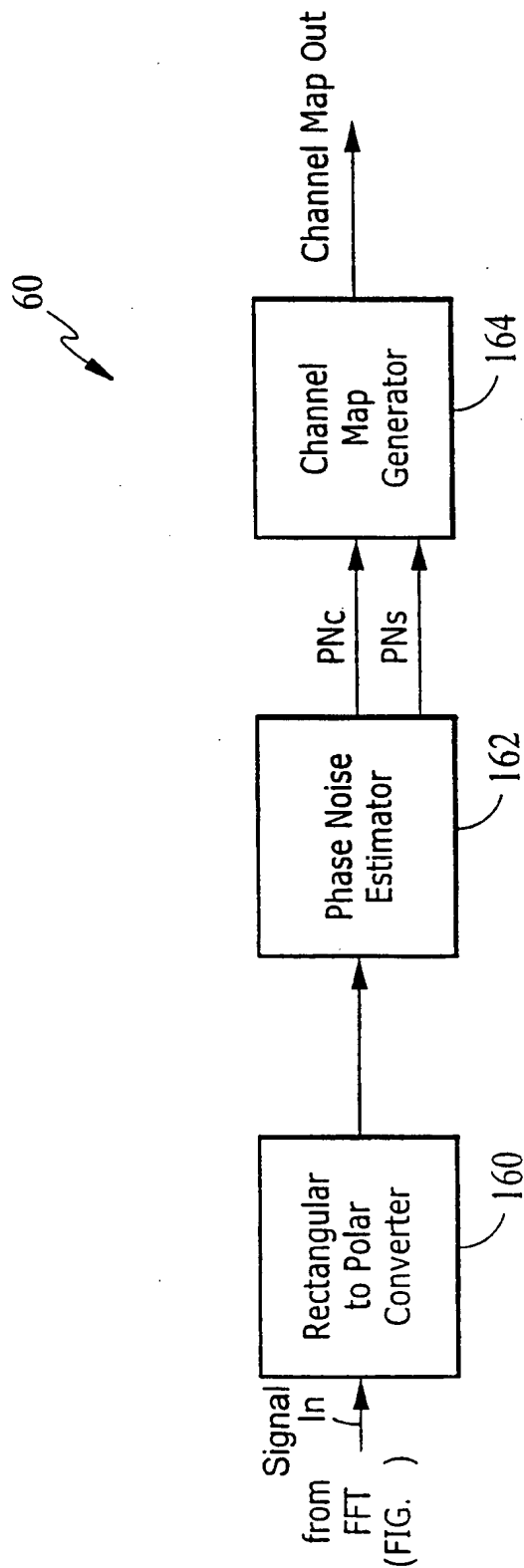


FIG. 8

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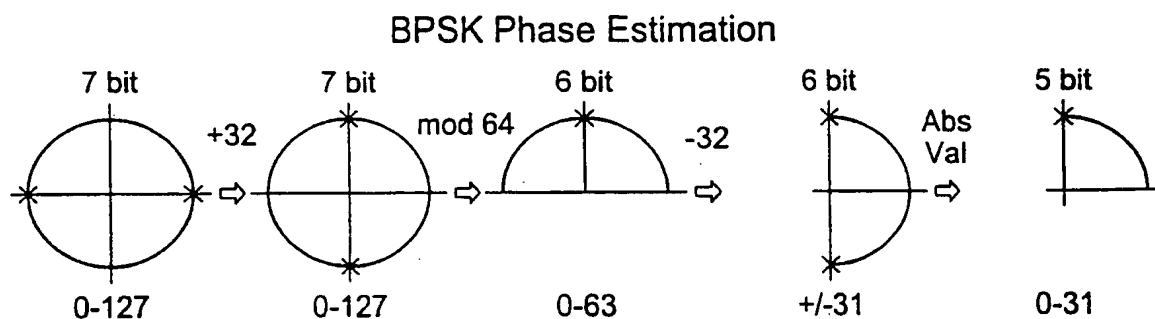


FIG. 9A

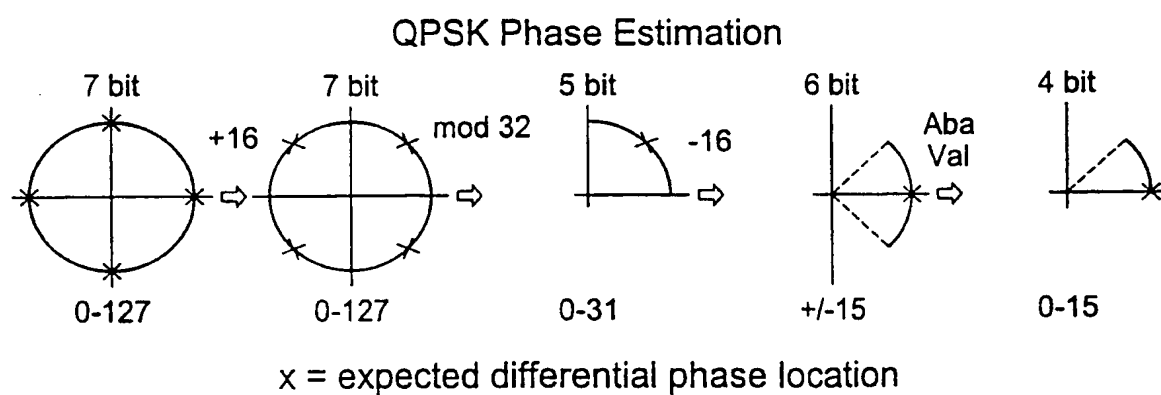


FIG. 9B

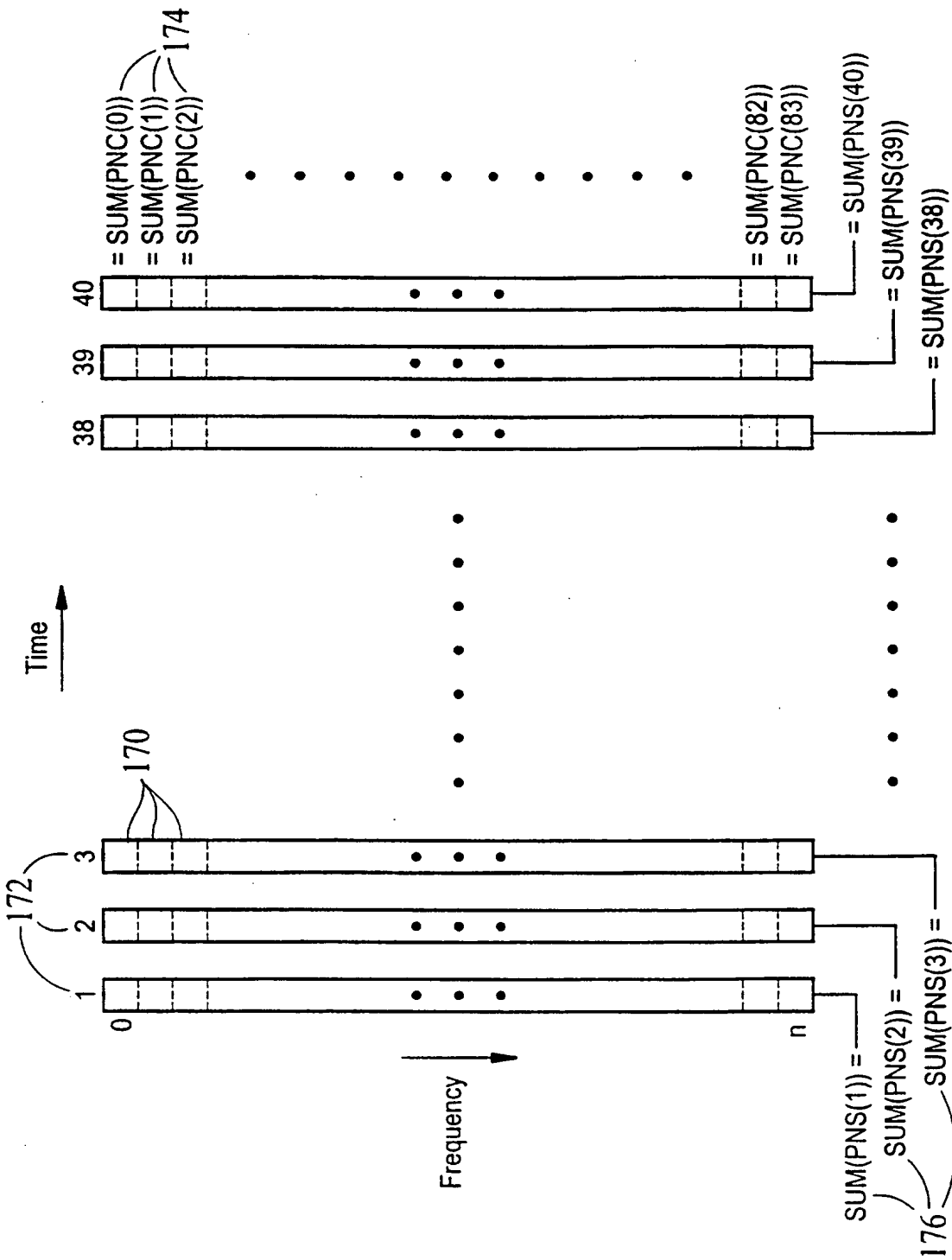


FIG. 10

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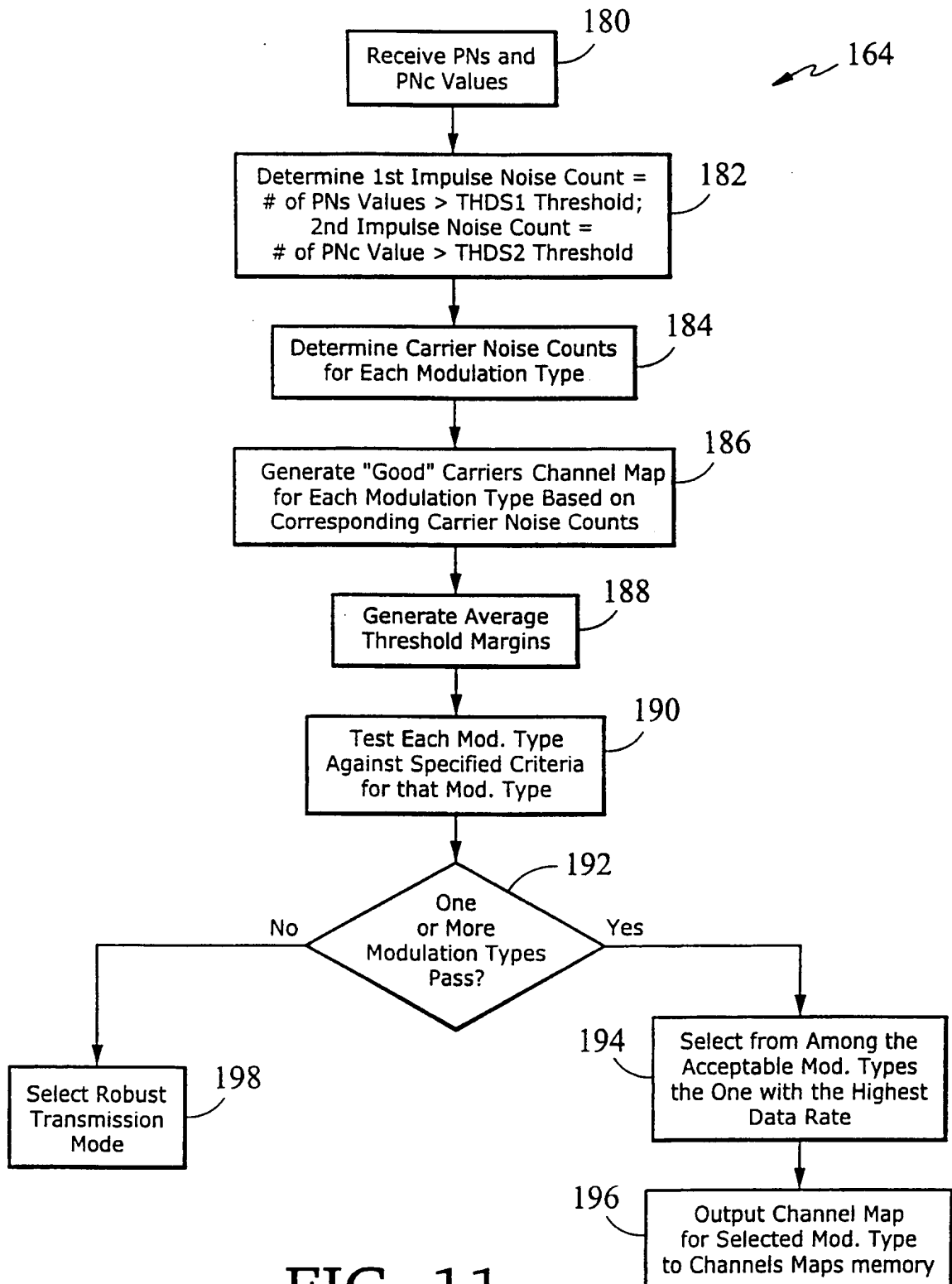


FIG. 11

INTERNATIONAL SEARCH REPORT

International application No.
PCT/US00/33090

A. CLASSIFICATION OF SUBJECT MATTER

IPC(7) : H/03/M 13/03; H/04/J 11/00, 3/26

US CL : 714/792; 370/203,206,208,432

According to International Patent Classification (IPC) or to both national classification and IPC

B. FIELDS SEARCHED

Minimum documentation searched (classification system followed by classification symbols)

U.S. : 714/792; 370/203,206,208,432

Documentation searched other than minimum documentation to the extent that such documents are included in the fields searched

Electronic data base consulted during the international search (name of data base and, where practicable, search terms used)

EAST, IEEE Website

C. DOCUMENTS CONSIDERED TO BE RELEVANT

Category*	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
X ---- Y	US 5963557 A (ENG) 05 October 1999, col. 9, lines 29-34; col. 4, lines 28-50; col. 12, lines 29-33; col. 11, lines 52-57; col. 5, lines 59-67; Figures 2 and 10A.	1-3, 13-16, 26-27 ----- 11, 12, 24, 25
Y, P	US 6158041 A (RALEIGH et al) 05 December 2000, col. 2, lines 19-21; Figures 1 and 2.	1-3, 11-16, 24-27
Y, P ---- A, P	US 6151296 A (VIJAYAN et al) 21 November 2000, col. 2, lines 64-67, col. 3, lines 1-8.	1-3, 13-16, 26, 27 ----- 4-12, 17-25
A	US 5966412 A (RAMASWAMY) 12 October 1999, col. 2, lines 20-27, Figure 1.	1-27

☒ Further documents are listed in the continuation of Box C.
 ☐ See patent family annex.

* Special categories of cited documents:	*T* later document published after the international filing date or priority date and not in conflict with the application but cited to understand the principle or theory underlying the invention
A document defining the general state of the art which is not considered to be of particular relevance	*X* document of particular relevance; the claimed invention cannot be considered novel or cannot be considered to involve an inventive step when the document is taken alone
E earlier document published on or after the international filing date	*Y* document of particular relevance; the claimed invention cannot be considered to involve an inventive step when the document is combined with one or more other such documents, such combination being obvious to a person skilled in the art
L document which may throw doubts on priority claim(s) or which is cited to establish the publication date of another citation or other special reason (as specified)	*G* document member of the same patent family
O document referring to an oral disclosure, use, exhibition or other means	
P document published prior to the international filing date but later than the priority date claimed	

Date of the actual completion of the international search 08 JANUARY 2001	Date of mailing of the international search report 09 APR 2001
Name and mailing address of the ISA/US Commissioner of Patents and Trademarks Box PCT Washington, D.C. 20231 Facsimile No. (703) 305-3230	Authorized officer <i>Joseph David Torres</i> JOSEPH DAVID TORRES Telephone No. (703) - 308 -7066

INTERNATIONAL SEARCH REPORT

International application No.
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C (Continuation). DOCUMENTS CONSIDERED TO BE RELEVANT

Category*	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
Y,P ----- A,P	US 6108353 A (NAKAMURA et al) 22 August 2000, col. 1, lines 11-19; col. 10, lines 31-58; Figures 4 and 5.	1-3, 13-16, 26, 27 ----- 4-12, 17-25
Y ----- A	US 5452288 A (RAHUEL et al) 19 September 1995, col. 3, lines 43-55, col. 26, lines 1-33; Figures 4 and 10.	1-3, 11-16, 24-27 ----- 4-10, 17-23
Y ----- A	US 5914932 A (SUZUKI et al) 22 June 1999, col. 2, lines 56-67; col. 10, lines 24-28; Figures 4 and 5.	1-3, 11-16, 24-27 ----- 4-10, 17-23
Y ----- A	US 5903614 A (SUZUKI et al) 11 May 1999, col. 2, lines 57-67; col. 4, lines 23-34; Figures 4, 5, 6, 7 and 8.	1-3, 11-16, 24-27 ----- 4-10, 17-23
Y ----- A	US 5228025 A (LE FLOCH et al) 13 July 1993, col. 2, lines 42-46; col. 9, lines 1-28; Figure 1.	1-3, 13-16, 26, 27 ----- 4-12, 17-25
Y ----- A	US 5694389 A (SEKI et al) 02 December 1997, col. 1, lines 44-63; col. 12, lines 50-67; Figures 3 and 4.	1-3, 11-16, 24-27 ----- 4-10, 17-23